

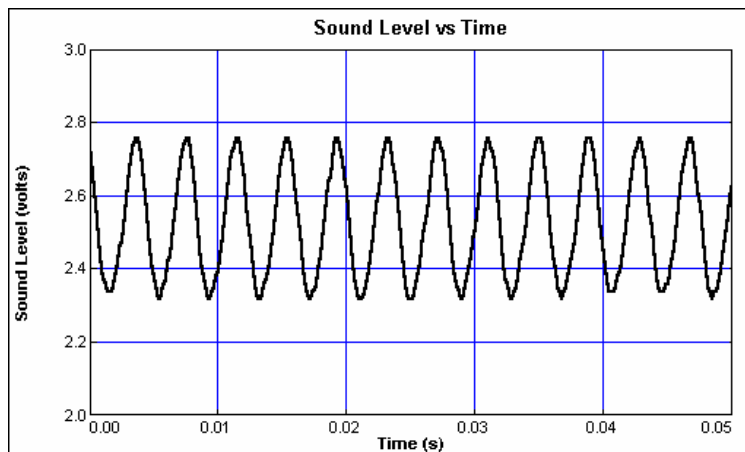
Sound Waves and Beats

Sound waves consist of a series of air pressure variations. A Microphone diaphragm records these variations by moving in response to the pressure changes. The diaphragm motion is then converted to an electrical signal. Using a Microphone and a computer interface, you can explore the properties of common sounds.

The first property you will measure is the *period*, or the time for one complete cycle of repetition. Since period is a time measurement, it is usually written as T . The reciprocal of the period ($1/T$) is called the *frequency*, f , the number of complete cycles per second. Frequency is measured in hertz (Hz). $1 \text{ Hz} = 1 \text{ s}^{-1}$.

A second property of sound is the *amplitude*. As the pressure varies, it goes above and below the average pressure in the room. The maximum variation above or below the pressure mid-point is called the amplitude. The amplitude of a sound is closely related to its loudness.

When two sound waves overlap, their air pressure variations will combine. For sound waves, this combination is additive. We say that sound follows the principle of *linear superposition*. Beats are an example of superposition. Two sounds of nearly the same frequency will create a distinctive variation of sound amplitude, which we call beats. You can study this phenomenon with a Microphone, lab interface, and computer.



OBJECTIVES

- Measure the frequency and period of sound waves from tuning forks.
- Measure the amplitude of sound waves from tuning forks.
- Observe beats between the sound of two tuning forks.

MATERIALS

Power Macintosh or Windows PC
LabPro or Universal Lab Interface
Vernier Microphone

Logger Pro
2 tuning forks or electronic keyboard



PRELIMINARY QUESTIONS

1. Why are instruments tuned before being played as a group? In which ways do musicians tune their instruments?
2. Given that sound waves consist of series of air pressure increases and decreases, what would happen if an air pressure increase from one sound wave was located at the same place and time as a pressure decrease from another of the same amplitude?

PROCEDURE

1. Connect the Vernier Microphone to Channel 1 of the LabPro or Universal Lab Interface.
2. Open the file in the Experiment 21 folder of *Physics with Computers*. The computer will take data for just 0.05 s to display the rapid pressure variations of sound waves. The vertical axis corresponds to the variation in air pressure and the units are arbitrary.


Part I Simple Waveforms

3. Produce a sound with a tuning fork or keyboard, hold it close to the Microphone and click . The data should be sinusoidal in form, similar to the sample on the front page of this lab. If you are using a tuning fork, strike it against a soft object such as a rubber mallet or the rubber sole of a shoe. Striking it against a hard object can damage it. If you strike the fork too hard or too softly, the waveform may be too rough; try again.
4. Note the appearance of the graph. Count and record the number of complete cycles shown after the first peak in your data.
5. Click the Examine button, . Drag the mouse across the graph and record the times for the first and last peaks of the waveform. Divide the difference, Δt , by the number of cycles to determine the period of the tuning fork.
6. Calculate the frequency of the tuning fork in Hz and record it in your data table.
7. Drag the mouse across the graph and record the maximum and minimum y values for an adjacent peak and trough.
8. Calculate the amplitude of the wave by taking half of the difference between the maximum and minimum y values. Record the values in your data table.
9. Make a sketch of your graph or print the graph.
10. Save your data by choosing Store Latest Run from the Data menu. Hide the run by choosing Hide Run ► Run 1 from the Data menu.
11. Repeat Steps 3 – 9 for the second frequency. Store the latest run. It will be stored as Run 2. Then hide Run 2.

Part II Beats

12. Two pure tones with different frequencies sounded at once will create the phenomenon known as beats. Sometimes the waves will reinforce one another and other times they will combine to a reduced intensity. This happens on a regular basis because of the fixed frequency of each tone. To observe beats, strike your tuning forks at the same time

(simultaneously) or simultaneously hold down two keys on the keyboard and listen for the combined sound. If the beats are slow enough, you should be able to hear a variation in intensity. If the beats are rapid a single rough-sounding tone is heard.

13. Collect data while the two tones are sounding. You should see a time variation of the sound amplitude. When using tuning forks, strike them equally hard and hold them the same distance from the Microphone. When you get a clear waveform, choose Store Latest Run from the Data menu. The beat waveform will be stored as Run 3.
14. The pattern will be complex, with a slower variation of amplitude on top of a more rapid variation. Ignoring the more rapid variation and concentrating in the overall pattern, count the number of amplitude maxima after the first maximum and record it in the data table.
15. Click the Examine button, . Drag the mouse across the graph and record the times for the first and last amplitude maxima. Divide the difference, Δt , by the number of cycles to determine the period of beats (in s). Calculate the *beat frequency* in Hz from the beat period. Record these values in your data table.

DATA TABLE

Part I Simple Waveforms

Tuning fork or note	Number of cycles	First maximum (s)	Last maximum (s)	Δt (s)	Period (s)	Calculated frequency (Hz)

Tuning fork or note	Peak (V)	Trough (V)	Amplitude (V)

Tuning fork or note	Parameter A (V)	Parameter B (s^{-1})	$f = B/2\pi$ (Hz)

Part II Beats

Number of cycles	First maximum (s)	Last maximum (s)	Δt (s)	Beat (s)	Calculated beat frequency (Hz)



ANALYSIS


Part I Simple Waveforms

1. In the following analysis, you will see how well a sine function model fits the data. The displacement of the particles in the medium carrying a periodic wave can be modeled with a sinusoidal function. Your textbook may have an expression resembling this one:

$$y = A \sin(2\pi f t)$$

In the case of sound, a longitudinal wave, the y refers to the change in air pressure that makes up the wave. A is the amplitude of the wave (a measure of loudness), and f is the frequency. Time is represented with t and the sine function requires a factor of 2π when evaluated in radians.

Logger *Pro* will fit the function $y = A * \sin(B * x + C) + D$ to experimental data. A , B , C , and D are parameters (numbers) that Logger *Pro* reports after a fit. This function is more complicated than the textbook model, but the basic sinusoidal form is the same. Comparing terms, listing the textbook model's terms first, the amplitude A corresponds to the fit term A , and $2\pi f$ corresponds to the parameter B . The time t is represented by the x , Logger *Pro*'s x -axis. The new parameters C and D shift the fitted function left-right and up-down, respectively and are necessary to obtain a good fit. Only the parameters A and B are important to this experiment. In particular, the numeric value of B allows you to find the frequency f using $B = 2\pi f$. Choose Show ► Run 1 from the Data menu to show the waveform from the first tone. Hide the other runs. Click the Curve Fit button, , select “ $A * \sin(B * x + C) + D$ ” from the list of models and “Sound Level Run 1” from the Perform Fit On list. Click  to perform the curve fit.

2. Click  to return to the graph. The model and its parameters appear in a floating box in the upper left corner of the graph. Record the parameters A and B of the model in your data table.
3. Since B corresponds to $2\pi f$ in the curve fit, use the curve fit information to determine the frequency. Enter the value in your data table. Compare this frequency to the frequency calculated earlier. Which would you expect to be more accurate? Why?
4. Compare the parameter A to the amplitude of the waveform.
5. Hide Run 1 and show Run 2, the waveform of the second tone. Repeat Steps 1 – 4 for Run 2.

Part II Beats

6. Is there any way the two individual frequencies can be combined to give the beat frequency you measured earlier? Compare your conclusion with information given in your textbook.